

A method and device for storing audio-centered information by a Table-of-Contents (TOC) mechanism and also by a file-based access mechanism through a ROOT directory that contains a highest level TOC directory, and a unitary storage medium containing such information.

The invention relates to a method as recited in the preamble of Claim 1.

Digital audio storage on unitary media such as disc or tape is at present widespread. If the audio is effectively sub-divided into multiple sub-items, the providing of a **Table-of-Contents (TOC)** allows to access the information in a relatively fast manner. Generally, a TOC file specifies at least **what** has been stored and **where** it has been stored, for facilitating access thereto in a home-type player. A lowest level TOC file, that in case of a single-level TOC structure is also the highest level TOC file, thereby points immediately to the content of the audio items or tracks. Now, such audio signals are also being used in the environment of personal computers and the like, where the scope of such use may have various extended features. An audio provider may wish to have such information straightforwardly accessible through various platforms as different as a portable player from a full-fledged multi-media engine.

In consequence, amongst other things, it is an object of the present invention to allow audio management on levels of different complexity, through providing parallel but compatible accessibility to players as well as to PC's, and in particular, through the latter's file structure. Now therefore, according to one of its aspects the invention is characterized according to the characterizing part of Claim 1. A lowest level directory, that in case of a single level directory structure is also the highest level or ROOT directory, contains localizing information of the associated audio items or tracks. However, such lowest level directory cannot access immediately the audio content of these tracks, but would need the appropriate application program to execute an interpretation with respect to the pertaining audio file.

Advantageous features of the invention are to provide a multi-level TOC structure, as well as a multi-level directory structure for raising access flexibility.

The invention also relates to a unitary storage medium produced by the method, and to a reader or player arranged for interfacing with such storage medium. Further advantageous aspects of the invention are recited in dependent Claims.

These and further aspects and advantages of the invention will be discussed

more in detail hereinafter with reference to the disclosure of preferred embodiments, and in particular with reference to the appended Figures that show:

Figures 1a, 1b a record carrier;

Figure 2 a playback device;

Figure 3 a recording device;

Figure 4, a file system for use with the invention;

Figure 5, a first storage arrangement for the invention;

Figure 6, a second storage arrangement for the invention.

Table 1 specifies a Master\_TOC Syntax;

Table 2 specifies a Master\_TOC\_O Syntax;

Table 3 specifies a Disc\_Info Syntax.

Figure 1a shows a disc-shaped record carrier 11 with track 19 and central hole 10. Track 19 is arranged in a spiral pattern of turns forming substantially parallel tracks on an information layer. The carrier may be an optical disc with a recordable or a prerecorded information layer. Examples of a recordable disc are CD-R, CD-RW, and DVD-RAM, whereas audio CD is a prerecorded disc. Prerecorded discs can be manufactured by first recording a master disc and later pressing consumer discs. Track 19 on the recordable record carrier is indicated by providing a pre-embossed track structure during manufacture of the blank record carrier. The track may be configured as a pregroove 14 to enable a read/write head to follow the track 19 during scanning. The information is recorded on the information layer by optically detectable marks along the track, e.g. pits and lands.

Figure 1b is a cross-section along the line b-b of a recordable record carrier 11, wherein transparent substrate 15 carries recording layer 16 and protective layer 17. The pregroove 14 may be implemented as an indentation, an elevation, or as a material property deviating from its surroundings.

For user convenience, the audio information on the record carrier has been subdivided into items, which usually have a duration of a few minutes e.g. songs on an album or movements of a symphony. Usually the record carrier also contains access information for identifying the items, such as in a so-called Table Of Contents (TOC), or included in a file system like ISO 9660 for CD-ROM. The access information may include playing time and start address for each item, and also further information like a song title.

The audio information is recorded in digital representation after analog to

digital (A/D) conversion. Examples of A/D conversion are PCM 16-bit per sample at 44.1 kHz known from CD audio and 1 bit Sigma Delta modulation at a high oversampling rate e.g. 64 x  $F_s$  called bitstream. The latter method represents a high quality encoding method, with the choice between high quality decoding and low quality decoding, the latter allowing a simpler decoding circuit. Reference is made in this respect to documents D5 and D6, *infra*. After A/D conversion, digital audio is compressed to variable bitrate audio data for recording on the information layer. The compressed audio data is read from the record carrier at such a speed, that after decompression substantially the original timescale will be restored when reproducing the audio information continuously. Hence the compressed data must be retrieved from the record carrier at a speed dependent on the varying bitrate. The data is retrieved from the record carrier at so-called transfer speed, i.e. the speed of transferring data bytes from the record carrier to a de-compressor. Preferably the record carrier has a constant spatial data density, which gives the highest data storage capacity per unit of area. In such system the transfer speed is proportional to the relative linear speed between the medium and the read/write head. If a buffer is provided before the de-compressor, actual transfer speed is the speed before that buffer.

Figure 2 shows a playback apparatus according to the invention for reading a record carrier 11 of the type shown in Figure 1. The device has drive means 21 for rotating record carrier 11 and a read head 22 for scanning the record carrier track. Positioning means 25 effect coarse radial positioning of read head 22. The read head comprises a known optical system with a radiation source for generating a beam 24 that is guided through optical elements and focused to spot 23 on an information layer track. The read head further comprises a focusing actuator for moving the focus of the radiation 24 along the optical axis of the beam and a tracking actuator for fine positioning of spot 23 in a radial direction on the centre of the track. The tracking actuator may comprise coils for moving an optical element or may be arranged for changing the angle of a reflecting element. The radiation reflected by the information layer is detected by a known detector in the read head 22, e.g. a four-quadrant diode, to generate a read signal and further detector signals including a tracking error and focusing error signals for the tracking and focusing actuators, respectively. The read signal is processed by a reading means 27 to retrieve the data, which reading means are of a usual type for example comprising a channel decoder and an error corrector. The retrieved data is passed to a data selection means 28, to select the compressed audio data for passing on to buffer 29. The selection is based on data type indicators also recorded on the record carrier, e.g. headers in a framed format. From buffer 29, the compressed audio data are passed on to de-compressor 31 as signal 30. This signal may also be outputted to an external de-compressor. De-compressor 31 decodes the compressed audio data to

reproduce the original audio information on output 32. The de-compressor may be fitted separately, e.g. in a stand-alone high quality audio digital to analog convertor (D/A convertor), as indicated by dashed rectangle 33 in Figure 2. Alternatively, the buffer may be positioned before the data selection means. The buffer 29 may be positioned in a separate housing or may be  
5 combined with a buffer in the decompressor. The device furthermore has a control unit 20 for receiving control commands from a user or from a host computer not shown, that via control lines 26 such as a system bus is connected to drive means 21, positioning means 25, reading means 27 and data selection means 28, and possibly also to buffer 29 for buffer filling level control. To this end, the control unit 20 may comprise control circuitry, such as a microprocessor,  
10 a program memory and control gates, for performing the procedures described below. Control unit 20 may be implemented as a logic circuit state machine.

The art of audio compression and de-compression is known. Audio may be compressed after digitizing by analyzing the correlation in the signal, and producing parameters for fragments of a specified size. During de-compression the inverse process is used to  
5 reconstruct the original signal. If the original digitized signal is reconstructed exactly, the (de-)compression is called lossless, whereas lossy (de)-compression will not reproduce certain details of the original signal which however are substantially undetectable by the human ear or eye. Most known systems for audio and video, such as DCC or MPEG, use lossy compression, whereas lossless compression is used for storing computer data. Examples of audio compression  
20 and decompression can be found in D2, D3 and D4 hereinafter, of which in particular the lossless compression from D2 is suitable for high quality audio.

According to the invention, data selection means 28 are arranged to retrieve from the data read certain control information. The data selection means 28 are also arranged to discard any padding or stuffing data, that had been added during recording. When the control unit  
25 20 is commanded to reproduce an item of audio from the record carrier, the positioning means 25 are controlled to position the reading head on the portion of the track containing the TOC. The starting address for that item will then be retrieved from the TOC via the data selection means 28. Alternatively the contents of the TOC may be read only once and stored in a memory when the disc is inserted in the apparatus. For reproducing the item, the drive means 21 are controlled  
30 to rotate the record carrier at the appropriate speed indicated through extracting time codes stored with the audio, that indicate an intended duration.

To provide continuous reproduction without buffer underflow or overflow the transfer speed is coupled to the reproduction speed of the D/A converter, i.e. to the bitrate after decompression. To this end the apparatus may comprise a reference frequency source for

controlling the decompressor. Alternatively or additionally the rotation rate may be adjusted using the average filling level of the buffer 29, e.g. decreasing the rotation rate when the buffer is more than 50% full on average.

Figure 3 shows a recording device for writing information on a record carrier according to the invention of a type which is (re)writable. During a writing operation, marks representing the information are formed on the record carrier. The marks may be in any optically readable form, e.g. in the form of areas whose reflection coefficient differs from their surroundings, through recording in materials such as dye, alloy or phase change, or in the form of areas with a direction of magnetization different from their surroundings when recording in magneto-optical material. Writing and reading of information for recording on optical disks and usable rules for formatting, error correcting and channel coding, are well-known, e.g. from the CD system. Marks may be formed through a spot 23 generated on the recording layer via a beam 24 of electromagnetic radiation from a laser diode. The recording device comprises similar basic elements as described with reference to Figure 2, i.e. a control unit 20, drive means 21 and positioning means 25, but it has a distinctive write head 39. Audio information is presented on the input of compression means 35, which may be placed in a separate housing. Suitable compression has been described in D2, D3 and D4. The variable bitrate compressed audio on the output of the compression means 35 is passed to buffer 36. From buffer 36 the data is passed to data combining means 37 for adding stuffing data and further control data. The total data stream is passed to writing means 38 for recording. Write head 39 is coupled to the writing means 38, which comprise for example a formatter, an error encoder and a channel encoder. The data presented to the input of writing means 38 is distributed over logical and physical sectors according to formatting and encoding rules and converted into a write signal for the write head 39. Unit 20 is arranged for controlling buffer 36, data combination means 37 and writing means 38 via control lines 26 and for performing the positioning procedure as described above for the reading apparatus. Alternatively the recording apparatus may be arranged for reading having the features of the playback apparatus and a combined write/read head.

Figure 4 shows a file system for use with the invention, for which in principle various options exist. As main choice the inventor has proposed that the storage medium should be based on either the UDF file system or the ISO 9660 file system, or both, which systems are by themselves standard to the skilled art person. In the alternative case, no file system should be present at all and the relevant sector spaces should be kept empty.

If a file system is present however, all audio will be stored in Audio Files.

Figure 4 has a double accessing hierarchy, as follows, Firstly, ROOT directory 50 points to

MASTER TOC files 52 and furthermore to subaltern directories 2C\_AUDIO 54, MC\_AUDIO 56, and PICTURES 66. The structure of **MASTER.TOC** 52 will be discussed hereinafter. Furthermore, **2C\_AUDIO** directory 54 points to TOC **2C\_AREA.TOC** 58 and in parallel therewith to the various stereo tracks **TRACKn.2CH** 60. Furthermore, there is **MC\_AUDIO** 5 directory 56. This points to TOC **MC\_AREA.TOC** 62 and in parallel therewith to the various stereo tracks **TRACKn.MCH** 64. In consequence, the tracks may be accessed either via the associated directories, or rather via the file system that may have **MASTER.TOC** and sub-TOCS as files. However, double storage facilities therefor are superfluous, inasmuch as the directories need only to **refer** to the TOC or TOCS. For reasons of safety, the master TOCs have been 10 provided in triplicate. Furthermore, the subaltern or AREA.TOCs have been provided in duplicate. Next, to audio, the system may be organized for containing still further items, such as pictures that also have their directory 66. As an additional extra, **2C\_TAREA.2CH** 59 and **MC\_TAREA.MCH** contain all associated tracks.

Figure 5 shows a first storage arrangement for use with the invention, which 5 by way of example has been represented as a single serial track. Along the horizontal axis the following items are evident. Item 120 is a **Lead-in** area that is used for mutually synchronizing the reader and the driving of the medium. Item 122 represents the **File System** that has been disclosed with reference to Figure 4. Item 124 represents a **TOC** that may be configured according to standard procedures and pertains to subsequent items **Stereo Audio Item** 126 and 20 **Multi-channel Audio Item** 128, and if necessary also to **Extra Data** Item 130. The size of item 124 need not be standardized, inasmuch as various different amounts of information may be present. Item 126 represents **Stereo Audio Information**, that may be defined according to a conventional standard and by itself does not constitute part of the invention. Item 128 represents **Multi Channel Audio Information**, that may be defined according to a conventional standard 25 and by itself does not constitute part of the invention. Generally, the two audio areas may have the same structure and contain the same piece of audio, apart from the distinguishing definitions of the various channels. The audio may be plain coded or lossless coded. All kinds of audio may be multiplexed with supplementary data, such as Compact Disc Text.

Item 130 represents **Extra Data Information** that may be defined in a 30 conventional standard and by itself does not form part of the invention. Item 132 represents a **Lead-Out Information**. The latter is in particular used during search operations. The lead-out may cover a ring of some 0.5 to 1 millimeter wide. According to the above, the stored information may be accessed either via the file system as laid down in item 122, or via the TOC structure laid down in item 124, and more particular, via a two- or multi-level TOC structure to

be discussed hereinafter.

Figure 6 shows a second storage arrangement for use with the invention, and in particular pertaining to a two-level TOC structure. Along the horizontal axis the following items are evident, next to items that have already been shown in Figure 3 and carry the same  
5 reference numerals. For clarity, items 120 and 132 have been suppressed.

Item 134 represents the **Master TOC** that begins at a uniformly standardized offset position with respect to the start of the Lead-in area at byte number **510**, copies at **520**, **530**. According to the embodiment, the Master-TOC measures only one standard-size sector and primarily contains pointers to the various Sub-TOCs or Area-TOCs to be disclosed hereinafter. A  
10 preferred syntax of the Master-TOC has a header with a **Signature** that identifies the Master-TOC, such as by "SACD Master TOC ". Further, Tables 1 and 2 specify the precise syntax of the MASTER\_TOC. The syntax has been given in elementary computer notation, together with the associated lengths and formats.

Master\_TOC\_Signature is an 8 byte string identifying the Master TOC. The value of  
15 Master\_TOC\_Signature must be "SACDMTOC" (\$53 \$41 \$43 \$44 \$4D \$54 \$4F \$43).

Likewise Table 3 specifies the disc info syntax in the same manner. In particular:

2CH\_TOC\_1\_Address is a 5 byte integer containing the logical sector number (LSN) of the first  
20 sector of Area TOC-1 in the 2 Channel Stereo Area. If the 2-Channel Stereo Area is not present, the value of 2CH\_TOC\_1\_Address must be zero.

2CH-TOC\_2\_Address is a 4 byte integer containing the LSN of the first sector of Area TOC-2 in the 2 Channel Stereo Area. if the 2-Channel Stereo Area is not present, the value of  
25 2CH\_TOC\_2\_Address must be zero.

MC\_TOC\_1\_Address is a 4 byte integer containing the LSN of the first sector of Area TOC-1 in the Multi Channel Stereo Area. If the Multi Channel Area is not present, the value of  
MC\_TOC\_1\_Address must be zero.

30 MC\_TOC\_2\_Address is a 4 byte integer containing the LSN of the first sector of Area TOC-2 in the Multi Channel Stereo Area. If the Multi Channel Area is not present, the value of  
MC\_TOC\_2\_Address must be zero.

The format of Disc\_Flags must be as follows: a hybrid bit and seven reserved bits.

The Hybr bit must be set to one on a Hybrid Disc. The Hybr bit must be set to zero on a not-Hybrid Disc.



## List of related documents

- (D1) Research Disclosure number 36411.  
August 1994, page 412-413
- (D2) PCT/IB97/01156 (PHN 16.452)
- 5 1 bit ADC and lossless compression of audio
- (D3) PCT/IB97/01303 (PHN 16.405)  
Audio compressor
- (D4) EP-A 402,973 (PHN 13.241)  
Audio compression
- 10 (D5) 'A digital decimating filter for analog-to-digital conversion of hi-fi audio signals',  
by J.J. van der Kam in Philips Techn. Rev. 42, no. 6/7, April 1986, pp. 230-8
- (D6) 'A higher order topology for interpolative modulators for oversampling A/D  
converters', by Kirk C.H. Chao et al in IEEE Trans. on Circuits and Systems, Vol  
37, no. 3, March 1990, pp. 309-18